

#3

09/576039 U.S. PRO  
05/23/00

367.38589X00

IN THE UNITED STATES PATENT AND TRADEMARK OFFICE

Applicant(s) : Andy PHELPS

Serial No. :

Filed: May 23, 2000

Title: AUDIO APPARATUS

Group:

LETTER CLAIMING RIGHT OF PRIORITY

Honorable Commissioner of  
Patents and Trademarks  
Washington, D.C. 20231

May 23, 2000

Sir:

Under the provisions of 35 USC 119 and 37 CFR 1.55, the applicant(s) hereby claim(s) the right of priority based on Great Britain Patent Application No.(s) 9913848.9 filed June 14, 1999.

A certified copy of said Great Britain Application is attached.

Respectfully submitted,

ANTONELLI, TERRY, STOUT & KRAUS, LLP

Carl I. Brundidge  
Registration No. 29,621

CIB/nac  
Attachment  
(703) 312-6600



The  
Patent  
Office



The Patent Office  
Concept House  
Cardiff Road  
Newport  
South Wales  
NP10 8QQ

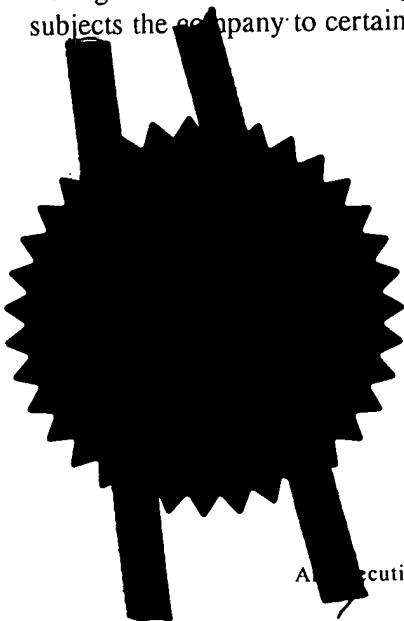
U.S. PRO  
09/576039  
05/23/00

I, the undersigned, being an officer duly authorised in accordance with Section 74(1) and (4) of the Deregulation and Contracting Out Act 1994, to sign and issue certificates on behalf of the Comptroller-General, hereby certify that annexed hereto is a true copy of the documents as originally filed in connection with the patent application identified therein together with the Statement of inventorship and of right to grant of a Patent (Form 7/77), which was subsequently filed.

In accordance with the Patents (Companies Re-registration) Rules 1982, if a company named in this certificate and any accompanying documents has re-registered under the Companies Act 1980 with the same name as that with which it was registered immediately before re-registration save for the substitution as, or inclusion as, the last part of the name of the words "public limited company" or their equivalents in Welsh, references to the name of the company in this certificate and any accompanying documents shall be treated as references to the name with which it is so re-registered.

In accordance with the rules, the words "public limited company" may be replaced by p.l.c., plc, P.L.C. or PLC.

Re-registration under the Companies Act does not constitute a new legal entity but merely subjects the company to certain additional company law rules.



Signed

Dated

18 FEB 2000

**CERTIFIED COPY OF  
PRIORITY DOCUMENT**

15JUN99 E454623-1 D02716  
P01/7700 0.00 - 9913848.9

## Request for grant of a patent

(See the notes on the back of this form. You can also get  
an explanatory leaflet from the Patent Office to help  
you fill in this form)

The Patent Office

Cardiff Road  
Newport  
Gwent NP9 1RH

1. Your reference

PAT 99009 GB

2. Patent application number

(The Patent Office will fill in this part)

14 JUN 1999

9913848.9

3. Full name, address and postcode of the or of  
each applicant (underline all surnames)NOKIA MOBILE PHONES LIMITED  
KEILALAHDENTIE 4  
02150 ESPOO  
FINLAND

Patents ADP number (if you know it) 7406747001 Rdes

If the applicant is a corporate body, give the  
country/state of its incorporation

FINLAND

4. Title of the invention

AUDIO APPARATUS

5. Name of your agent (if you have one)

"Address for service" in the United Kingdom  
to which all correspondence should be sent  
(including the postcode)NOKIA IPR DEPARTMENT  
NOKIA HOUSE  
SUMMIT AVENUE  
FARNBOROUGH  
HAMPSHIRE  
GU140NG UK

7577638001 ✓

Patents ADP number (if you know it)

6. If you are declaring priority from one or more  
earlier patent applications, give the country  
and the date of filing of the or of each of these  
earlier applications and (if you know it) the or  
each application numberCountry      Priority application number  
(if you know it)      Date of filing  
(day / month / year)7. If this application is divided or otherwise  
derived from an earlier UK application,  
give the number and the filing date of  
the earlier applicationNumber of earlier application      Date of filing  
(day / month / year)8. Is a statement of inventorship and of right  
to grant of a patent required in support of  
this request? (Answer 'Yes' if:

- a) any applicant named in part 3 is not an inventor, or
- b) there is an inventor who is not named as an  
applicant, or
- c) any named applicant is a corporate body.

See note (d)

Patents Form 1/77

9. Enter the number of sheets for any of the following items you are filing with this form.  
Do not count copies of the same document

Continuation sheets of this form

Description	9	/
Claim(s)	2	/
Abstract	1	/
Drawing(s)	4	+ 4

10. If you are also filing any of the following, state how many against each item.

Priority documents

Translations of priority documents

Statement of inventorship and right to grant of a patent (Patents Form 7/77)

Request for preliminary examination and search (Patents Form 9/77)

1 / ✓

Request for substantive examination (Patents Form 10/77)

Any other documents  
(please specify)

11.

I/We request the grant of a patent on the basis of this application.

Signature

JULIET HIBBERT

Date 14/06/99

12. Name and daytime telephone number of person to contact in the United Kingdom

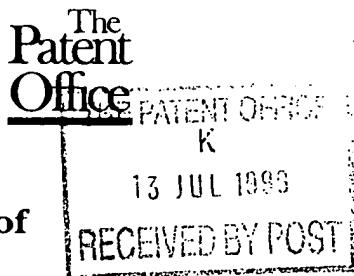
CHRIS HARRISON 01252 866000

**Warning**

After an application for a patent has been filed, the Comptroller of the Patent Office will consider whether publication or communication of the invention should be prohibited or restricted under Section 22 of the Patents Act 1977. You will be informed if it is necessary to prohibit or restrict your invention in this way. Furthermore, if you live in the United Kingdom, Section 23 of the Patents Act 1977 stops you from applying for a patent abroad without first getting written permission from the Patent Office unless an application has been filed at least 6 weeks beforehand in the United Kingdom for a patent for the same invention and either no direction prohibiting publication or communication has been given, or any such direction has been revoked.

**Notes**

- a) If you need help to fill in this form or you have any questions, please contact the Patent Office on 0645 500505.
- b) Write your answers in capital letters using black ink or you may type them.
- c) If there is not enough space for all the relevant details on any part of this form, please continue on a separate sheet of paper and write "see continuation sheet" in the relevant part(s). Any continuation sheet should be attached to this form.
- d) If you have answered 'Yes' Patents Form 7/77 will need to be filed.
- e) Once you have filled in the form you must remember to sign and date it.
- f) For details of the fee and ways to pay please contact the Patent Office.



**Statement of inventorship and of  
right to grant of a patent**

The Patent Office  
Cardiff Road  
Newport  
Gwent NP9 1RH

## 1. Your reference

PAT 99009 GB

2. Patent application number  
(if you know it)

9913848.9

## 3. Full name of the or of each applicant

NOKIA MOBILE PHONES LIMITED  
KEILALAHDENTIE 4  
02150 ESPOO  
FINLAND

## 4. Title of the invention

AUDIO APPARATUS

5. State how the applicant(s) derived the right  
from the inventor(s) to be granted a patent

BY VIRTUE OF SECTION 7(2)(c).

6. How many, if any, additional Patents Forms  
7/77 are attached to this form?  
(see note (c))

I/We believe that the person(s) named over the page (and on  
any extra copies of this form) is/are the inventor(s) of the invention  
which the above patent application relates to.

Signature  Date 9/7/99

JULIET JANE GRACE HIBBERT

8. Name and daytime telephone number of  
person to contact in the United Kingdom

CHRIS HARRISON 01252 865037

## Notes

- a) If you need help to fill in this form or you have any questions, please contact the Patent Office on 0645 500505.
- b) Write your answers in capital letters using black ink or you may type them.
- c) If there are more than three inventors, please write the names and addresses of the other inventors on the back of another Patents Form 7/77 and attach it to this form.
- d) When an application does not declare any priority, or declares priority from an earlier UK application, you must provide enough copies of this form so that the Patent Office can send one to each inventor who is not an applicant.
- e) Once you have filled in the form you must remember to sign and date it.

Enter the full names, addresses and postcodes of the inventors in the boxes and underline the surnames

ANDREW PHELPS  
103 MALMESBURY ROAD  
SOUTHHAMPTON  
HAMPSHIRE  
SO15 5FP

7698707001

Patents ADP number (*if you know it*):

Patents ADP number (*if you know it*):

**Reminder**

Have you signed the form?

Patents ADP number (*if you know it*):

**AUDIO APPARATUS**

5 This invention relates to audio apparatus, and in particular audio apparatus for radiating an ultrasonic pressure wave with a modulated audio signal.

10 Audio spotlighting, using a parametric audio system, provides a means for projecting highly directional beams of audible sound. This technology employs the non-linearity of a compressible material (such as air) to create audible by-products from inaudible ultrasound. This technique exploits an acoustic phenomenon called self-demodulation where low frequency audio beams of high directivity are generated from a high amplitude ultrasonic beam that has been modulated with an audio signal. Self-demodulation generates new 15 frequencies in the received sound spectra based on the envelope frequency (i.e. the required audio signal) in a process akin to AM demodulation. Thus this technique provides a beam of audio sound with the focused directional properties of the original ultrasonic carrier beam, allowing distant targeting of specific listeners. This technique can produce predictable and controllable 20 levels of audio sound and, despite the relatively weak effect of self-demodulation, is capable of generating substantial sound amplitudes due to the narrow spatial distribution of the acoustic energy.

25 The transmitted modulated ultrasonic wave can be considered as a collimated primary wave consisting of an amplitude modulated wave of pressure where the primary pressure wave is defined by

$$1) \quad p_1 = P_1 E(t) \sin(\omega_c t)$$

where  $P_1$  is the amplitude of the primary beam pressure,  $\omega_c$  is the carrier frequency and  $E(t)$  is the modulation envelope. For an amplitude modulated signal  $E(t)$  is equal to  $(1 + mg(t))$  where  $m$  is the modulation depth and  $g(t)$  is the audio signal.

5

As a result of  $p_1$  interacting with the air the modulated audio signal demodulates creating an audible secondary pressure wave  $p_2$  given by

$$10 \quad 2) \quad p_2(t) = \frac{\beta P_1^2 A}{16\pi\rho_0 c_0^4 z \alpha} \frac{\partial^2}{\partial t^2} E^2(\tau)$$

where  $\beta$  is the coefficient of nonlinearity ( $\beta_{\text{air}} = 1.2$ ),  $\rho_0$  is the ambient density of the medium,  $c_0$  is the small signal wave propagation speed,  $A$  is the beam's cross-sectional area,  $z$  is the axial distance,  $\alpha$  is the absorption coefficient of the medium. So, for example, where  $c=343$  m/s,  $\rho_0 = 1.2\text{kg/m}^3$ ,  $\alpha = 0.6$ , and  $A=5\times10\text{m}^{-3}$ , a 140dB ultrasound wave modulated with a 1kHz signal would produce about 71dB of audible sound at 1m.

20 The power of the resultant audio signal is proportional to the second derivative of the square of the modulation envelope. As a result significant coloration (i.e. a shift of signal power with respect to frequency) and distortion are introduced onto the demodulated audio signal as a result of the interaction of the ultrasonic wave with the non-linear medium. The coloration of the signal results in the low frequency audio components being suppressed by 25 approximately 12dB/octave; this is represented by the second derivative term of the modulation envelope. The distortion of the signal is represented by the square of the modulation envelope.

30 Processing the audio signal prior to modulation can minimise the effects of coloration and distortion that result from the interaction of the ultrasonic wave

with the non-linear medium. The processing typically comprises a double integration filter to compensate for coloration of the audio signal and a square root operation to compensate for the distortion of the audio signal.

5 However, for the self-demodulation to occur high ultrasonic sound pressure levels are required. To generate these high pressure levels it is necessary to generate the ultrasonic pressure levels at or close to the resonant frequency of the transmitting transducer. Correspondingly the frequency response of the transducer can vary dramatically at this frequency. The variable transducer  
10 frequency response can significantly affect the quality of the demodulated audio pressure wave.

Figure 1, plot A shows the frequency spectrum of a white noise input signal constrained between 300 and 4000 Hz prior to modulation with an ultrasonic  
15 carrier signal. An example of the effects of self-demodulation and transducer conversion upon the input signal, using a typical transducer having a measured frequency response shown in figure 2, is shown in figure 1, plot B.

The frequency response of a transducer can be flattened at the resonant  
20 frequency. However this requires considerable damping to be added to the transducer, and a corresponding drop in ultrasonic pressure level. This in turn would require a transducer with a large radiating surface area, which is not suitable for small devices, for example a mobile communication device and in particular a radiotelephone.

25

In accordance with an aspect of the present invention there is provided audio apparatus comprising a modulator for modulating a first ultrasonic signal with an audio signal to provide a second ultrasonic signal; a transducer for converting the second ultrasonic signal into an ultrasonic pressure wave for  
30 transmission into a non-linear medium to allow demodulation of the ultrasonic

pressure wave and thereby generate an audio pressure wave representative of the audio signal; processing means for modifying the audio signal to compensate for the demodulating properties of the non-linear medium; and means for modifying the audio signal to compensate for the conversion 5 characteristics of the transducer.

This provides the advantage of enabling the effects of the demodulation process and the transducer conversion characteristics on the audio signal to be minimised. This can allow the size of the transducer to be reduced while 10 retaining the performance of the transducer.

Typically the means for modifying the audio signal to compensate for the conversion characteristics of a transducer is a transducer response filter.

15 Most preferably the processing means comprises a double integration filter and a square root operator. As the operator processes the audio signal non-linearly the characteristics of the second filter are preferably derived empirically by tone adjustment for the required frequency range of the audio signal.

20

The invention will now be described, by way of one example only, with reference to the accompanying drawings, in which:

Figure 1, plot A shows the frequency spectrum of a test audio input signal to 25 an audio apparatus according to one embodiment of the present invention;

Figure 1, plot B shows the frequency spectrum of figure 1, plot A after self-demodulation without correction;

Figure 2 shows a typical measured frequency response of a suitable transducer for use in audio apparatus according to an embodiment of the present invention;

5 Figure 3 shows a radiotelephone having audio apparatus according to an embodiment of the present invention;

Figure 4 shows audio apparatus according to an embodiment of the present invention;

10

Figure 5, plot A shows the frequency spectrum of the test audio signal output after self-demodulation with correction for self-demodulation;

15 Figure 5, plot B shows the frequency spectrum of the test audio signal output from audio apparatus according to an embodiment of the present invention.

Figure 3 shows a radiotelephone 1 with audio apparatus 2, a speech decoder 3, a channel decoder 4, a receiver 5 and an antenna 6. In operation the receiver 5 receives a speech encoded digital signal 20 from a base station 20 (not shown) via antenna 6. The receiver 5 demodulates the received digital signal 20 and passes the demodulated signal 21 to channel decoder 4 which corrects for errors that may have occurred during the transmission process by using error protection bits encoded within the received signal. The receiver 5 typically samples the received signal 20 at 8 kHz. The decoded digital signal 25 22 is provided to speech decoder 3 which decodes the speech and passes the digital decoded signal 23 to audio apparatus 2 to generate an acoustic representation of the received speech signal as described in detail below. The audio apparatus 2 may be mounted within the radiotelephone 1. Typically however, to obtain the required audio power levels and to support hands free 30 use of the radiotelephone 1, the audio apparatus 2 is mounted separately to

the radiotelephone 1, for example, in conjunction with a hands free kit or a car kit for hands free use of a radiotelephone in a car.

Figure 4 shows the audio apparatus 2 which comprises a double integration filter 7, a transducer response filter 8, a DC shifter 9, an up-sampler 10, a square root operator 11, a modulator 12, an ultrasonic signal source 13, a digital to analogue converter 14 and an ultrasonic transducer 15.

The audio apparatus 2 is a parametric device that radiates an inaudible ultrasonic pressure wave with an audio signal modulated onto the ultrasonic pressure wave. The transmitted ultrasonic wave interacts with air (i.e. a compressible non-linear medium) to cause the ultrasonic wave to self-demodulate, thereby causing the modulated audio signal to become audible.

Digital signal 23 is provided to double integration filter 7, which boosts the low frequencies by 12 dB/octave akin to integrating the signal twice. Double integration filter 7 compensates for the effects of coloration that occur during the self-demodulation process, and is linear in nature. Typically, the double integration filter 7 is a simple recursive filter. The double integration filter 7 provides the double integrated digital signal 24 to transducer response filter 8.

The transducer response filter 8 corrects for characteristics of the ultrasonic transducer 15, as described in detail below. The transducer response filter 8 provides the corrected signal 25 to DC shifter 9.

25

The DC shifter 9 re-scales the data and shifts the voltage of the corrected digital signal 25 so that all signal voltages are positive, thus ensuring the square root operation only has to work on positive values, thereby avoiding complex filtering.

30

The DC shifted signal 26 is provided to up-sampler 10. The up-sampler 10 re-samples the 8 kHz signal at typically 120 kHz. The purpose of up-sampler 10 is to increase the frequency range of the signal in preparation for the square rooting of the signal. A consequence of square rooting the received signal is

- 5 the creation of an infinite series of harmonics. For distortion to be eliminated all these harmonics must be reproduced. Therefore, to ensure harmonics above 4 kHz are reproduced the signal is re-sampled at a higher frequency. Sample rates other than 120 kHz may be used dependant on the operating frequencies of the ultrasonic transducer. The re-sampled signal 27 is provided
- 10 to the square root operator 11.

The square rooting operator 11 compensates for the effects of distortion that occur during the self-demodulation process, and is non-linear in nature.

- 15 The square root operator 11 is typically performed by means of a look up table, as is well known to a person skilled in the art.

The square rooted signal 28 is provided to modulator 12 for modulation with an ultrasonic signal 29 from ultrasonic signal source 13. To minimise the risk

- 20 of harm to humans or animals the ultrasonic frequency should be higher than approximately 40kHz. Due to increased signal absorption by the air at higher frequencies the upper highest feasible frequency limit is typically of the order of 200kHz.

- 25 The digital modulated ultrasonic signal 30 is provided to digital to analogue converter 14 for converting the digital signal 30 to a representative analogue signal. The analogue modulated ultrasonic signal 31 is provided to ultrasonic transducer 15. Transducer 15 radiates the modulated ultrasonic signal as an inaudible ultrasonic pressure wave.

To obtain the required ultrasonic pressure levels required for self-demodulation to occur the transducer 15 will typically be chosen to have its resonance frequency at the frequency of the ultrasonic carrier signal 29.

5 For a 40 kHz ultrasonic signal a suitable transducer is the MuRata MA40B8S. This transducer has a frequency response as shown in figure 2, which has a narrow resonance band at 40 kHz. To obtain the required power levels a plurality of transducers will typically be required, for example 19 transducers will provide 55 dB's of audio speech signal.

10

As the square root operator 11 is non-linear it is not possible for the transducer response filter 8 to have a single optimum filtering characteristic for all frequencies. Therefore, the transducer response filter 8 is determined empirically for a required frequency range. For example, if the resultant

15 spectra of the self-demodulated signal shows a gradual power drop from 300 Hz to 4 kHz after correction for self-demodulation, the transducer response filter 8 is selected to boost the signal over this frequency range. The characteristics of the transducer response filter are therefore, typically, the inverse response of the resultant spectra of the self-demodulated signal. The

20 transducer response filter is designed typically using a recursive filter design package, for example the Yule-Walk package, which models the transducer response filter characteristics using the inverse of the self-demodulated signal.

25 Typically the transducer response filter characteristics are determined during the manufacture of the audio device. However, it is possible for the characteristics of the transducer response filter to be determined dynamically, for example while during use of the audio device.

Figure 5, plot A shows the transducer conversion effect upon the input signal shown in figure 1, plot A. Figure 5, plot B shows the corrected spectrum using a transducer response filter 8 empirically derived using figure 5, plot A for determining the effect of the transducer conversion. For different frequency 5 and modulation depths the square root non-linear operator 11 will vary the effects of the transducer conversion and accordingly different frequency and modulation depths will require the transducer response filter 8 to be modified empirically as described above.

10. The transducer response filter 8 is typically a simple recursive filter.

The present invention may include any novel feature or combination of features disclosed herein either explicitly or implicitly or any generalisation thereof irrespective of whether or not it relates to the presently claimed 15 invention or mitigates any or all of the problems addressed. In view of the foregoing description it will be evident to a person skilled in the art that various modifications may be made within the scope of the invention. The applicant hereby gives notice that new claims may be formulated to such features during prosecution of this application or of any such further application derived 20 therefrom. For example, it will be appreciated that an analogue audio signal can be processed and modulated with an ultrasonic carrier signal, also that the pre-processing filters and/or operator can be used to process the audio signal after modulation of the audio signal.

**CLAIMS**

1. Audio apparatus comprising a modulator for modulating a first ultrasonic signal with an audio signal to provide a second ultrasonic signal; a transducer for converting the second ultrasonic signal into an ultrasonic pressure wave for transmission into a non-linear medium to allow demodulation of the ultrasonic pressure wave and thereby generate an audio pressure wave representative of the audio signal; processing means for modifying the audio signal to compensate for the demodulating properties of the non-linear medium; and means for modifying the audio signal to compensate for the conversion characteristics of the transducer.
2. Audio apparatus according to claim 1, wherein the first ultrasonic signal is amplitude modulated with the audio signal.
3. Audio apparatus according to claim 1 or 2, wherein the first ultrasonic signal is equal to or greater than 40 kHz.
- 20 4. Audio apparatus according to any preceding claim, wherein the processing means comprises a double integration filter and a square root operator.
5. Audio apparatus according to claim 4, wherein the means for modifying is disposed between the double integration filter and the square root operator.
- 25 6. Audio apparatus according to any of the preceding claims, wherein the means for modifying is a digital filter.

7. Audio apparatus according to any preceding claim, wherein the characteristics of the means for modifying are empirically derived by tone adjustment.
- 5 8. A radiotelephone having audio apparatus according to any of the preceding claims.
9. A portable radio device having audio apparatus according to any of the preceding claims.
- 10 10. Audio apparatus substantially as hereinbefore described with reference to the accompanying drawings, and/or as shown therein.

**ABSTRACT**AUDIO APPARATUS

- 5    Audio apparatus comprising a modulator 12 for modulating a first ultrasonic signal 29 with an audio signal to provide a second ultrasonic signal 30; a transducer 15 for converting the second ultrasonic signal 30 into an ultrasonic pressure wave for transmission into a non-linear medium to allow demodulation of the ultrasonic pressure wave and thereby generate an audio
- 10    pressure wave representative of the audio signal 28; processing means 11 for modifying the audio signal to compensate for the demodulating properties of the non-linear medium; and means 8 for modifying the audio signal to compensate for the conversion characteristics of the transducer 15.

15

(Figure 4 )

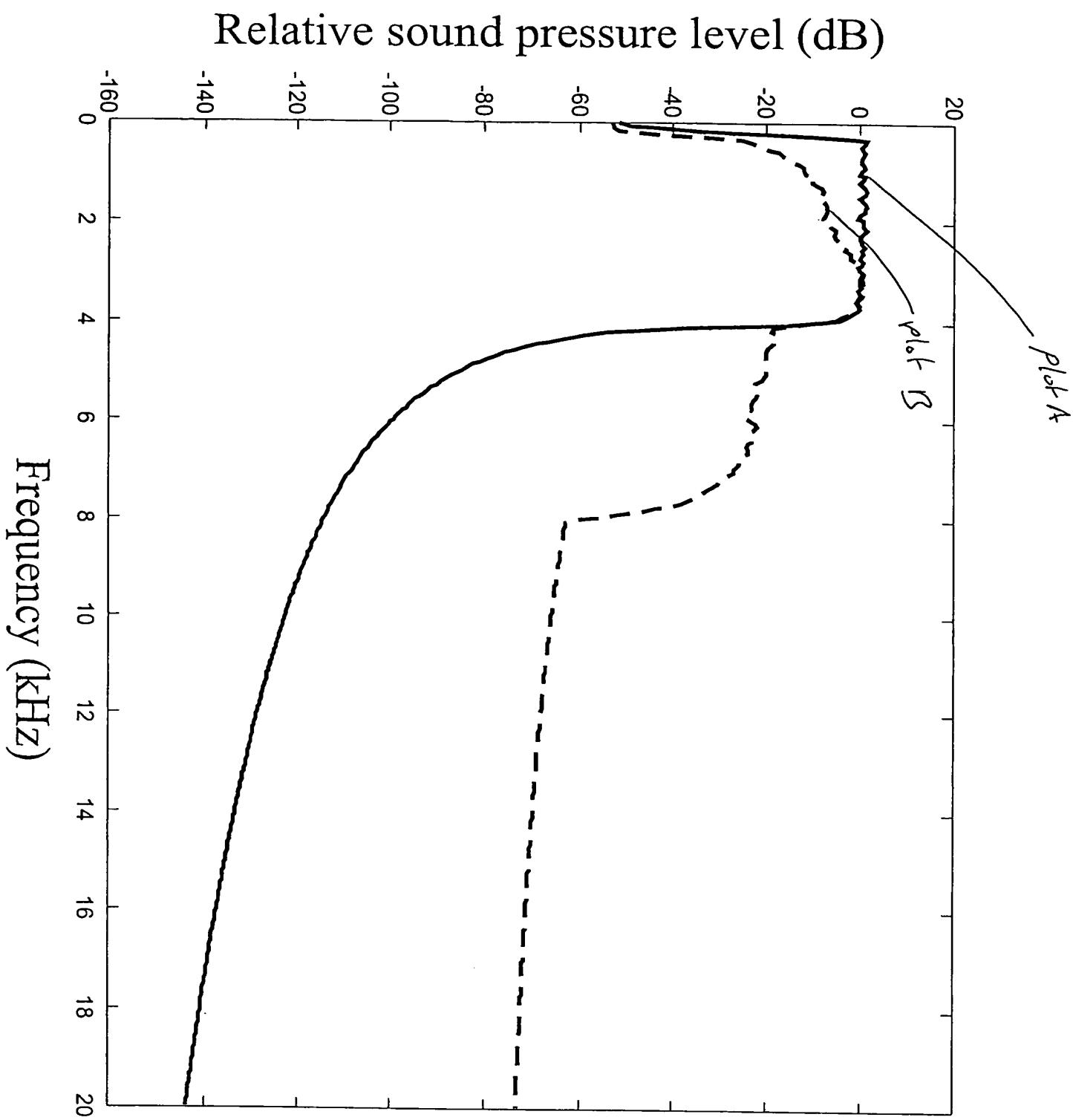


Figure 1

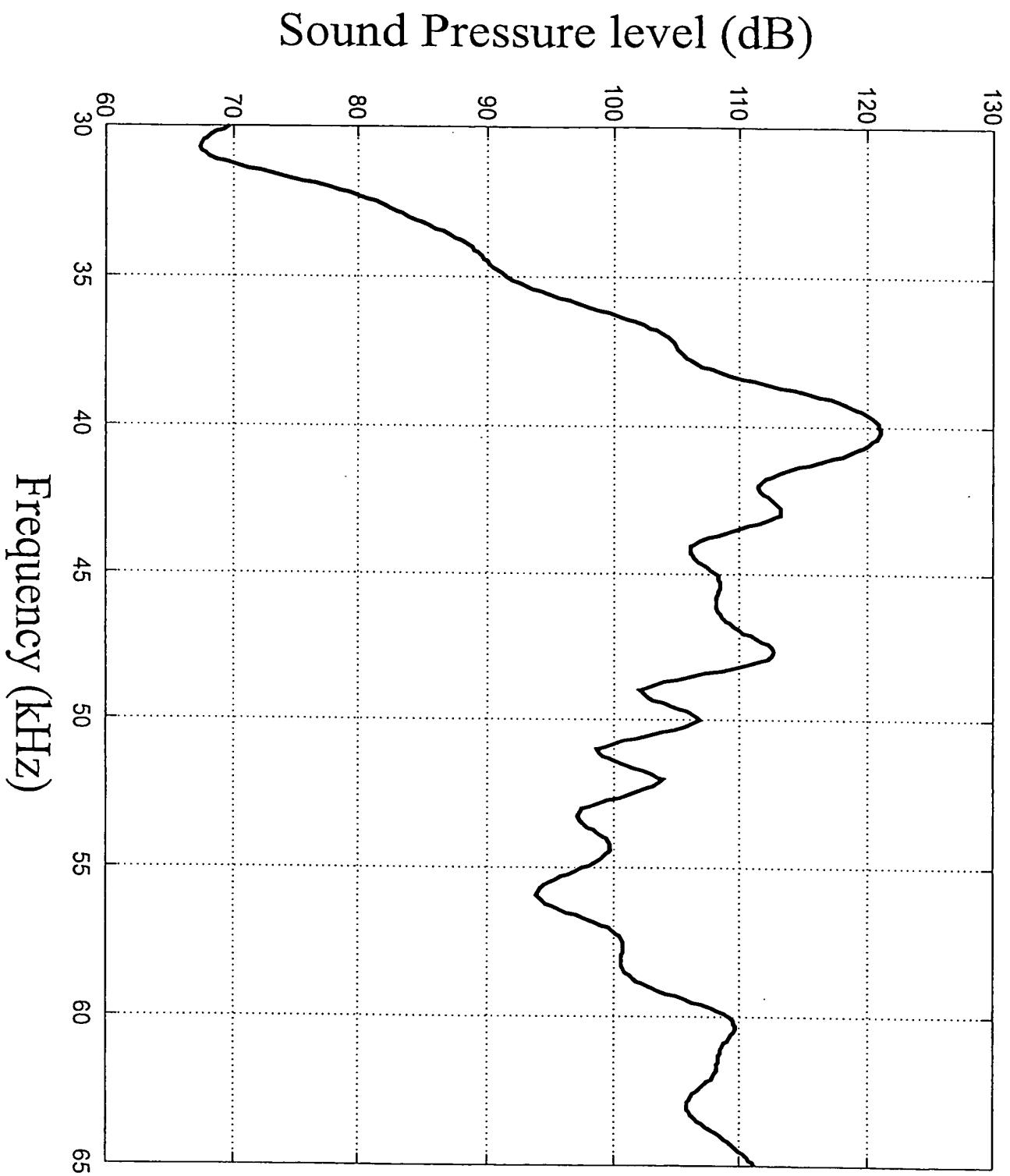


Figure 2

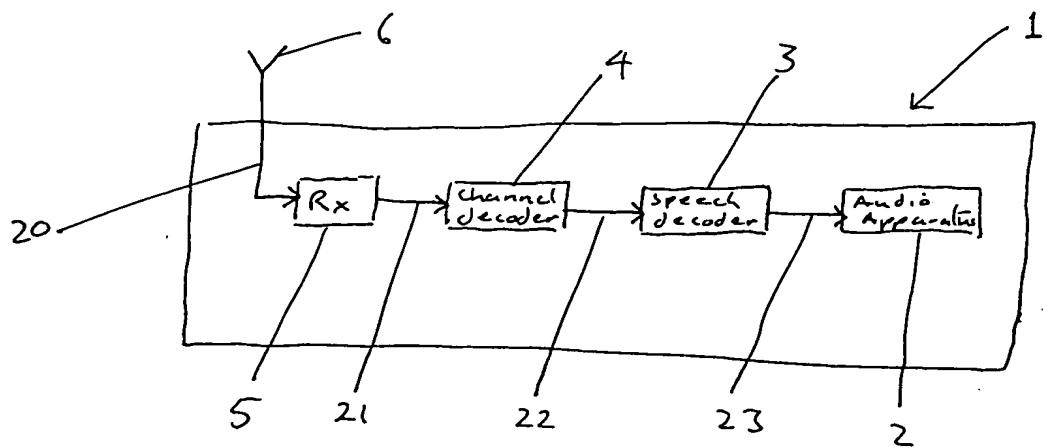


Figure 3

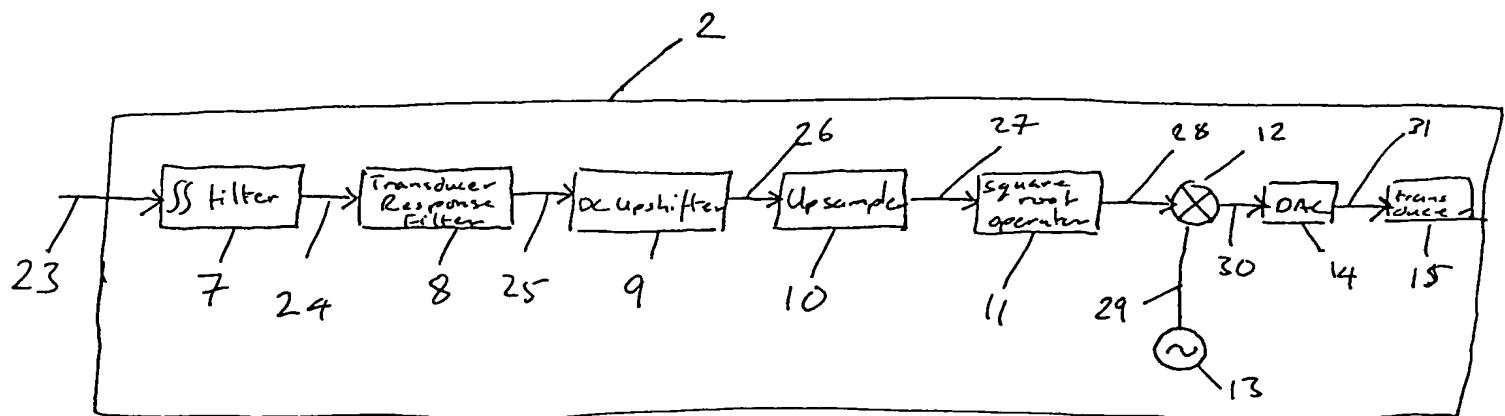


Figure 4

